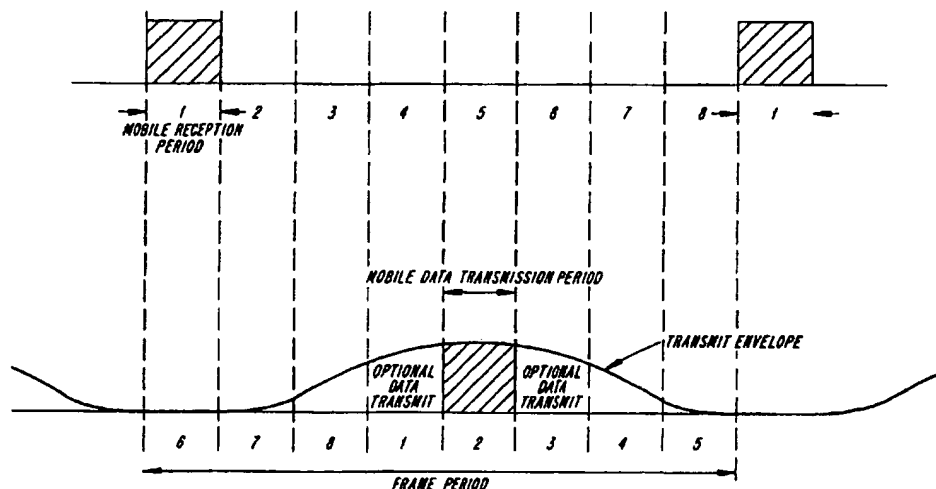


PCTWORLD INTELLECTUAL PROPERTY ORGANIZATION
International Bureau

INTERNATIONAL APPLICATION PUBLISHED UNDER THE PATENT COOPERATION TREATY (PCT)

(51) International Patent Classification ⁶ : H04Q 7/22, H04B 15/00	A1	(11) International Publication Number: WO 97/27712 (43) International Publication Date: 31 July 1997 (31.07.97)
(21) International Application Number: PCT/US97/00711 (22) International Filing Date: 17 January 1997 (17.01.97) (30) Priority Data: 08/590,891 24 January 1996 (24.01.96) US (71) Applicant: ERICSSON INC. [US/US]; 7001 Development Drive, P.O. Box 13969, Research Triangle Park, NC 27709 (US). (72) Inventors: DENT, Paul, Wilkinson; 637 Eaglepoint Road, Pittsboro, NC 27312 (US). TORSTENSSON, Anders; 103 Bastille Court, Cary, NC 27511 (US). (74) Agents: GRUDZIECKI, Ronald, L. et al.; Burns, Doane, Swecker & Mathis, L.L.P., P.O. Box 1404, Alexandria, VA 22313-1404 (US).		(81) Designated States: AL, AM, AT, AU, AZ, BA, BB, BG, BR, BY, CA, CH, CN, CU, CZ, DE, DK, EE, ES, FI, GB, GE, HU, IL, IS, JP, KE, KG, KP, KR, KZ, LC, LK, LR, LS, LT, LU, LV, MD, MG, MK, MN, MW, MX, NO, NZ, PL, PT, RO, RU, SD, SE, SG, SI, SK, TJ, TM, TR, TT, UA, UG, UZ, VN, ARIPO patent (KE, LS, MW, SD, SZ, UG), Eurasian patent (AM, AZ, BY, KG, KZ, MD, RU, TJ, TM), European patent (AT, BE, CH, DE, DK, ES, FI, FR, GB, GR, IE, IT, LU, MC, NL, PT, SE), OAPI patent (BF, BJ, CF, CG, CI, CM, GA, GN, ML, MR, NE, SN, TD, TG). Published <i>With international search report.</i> <i>Before the expiration of the time limit for amending the claims and to be republished in the event of the receipt of amendments.</i>

(54) Title: CORDLESS TIME-DUPLEX PHONE WITH IMPROVED HEARING-AID COMPATIBLE MODE**(57) Abstract**

Techniques for providing a hearing aid compatible mode of transmission for a mobile station in a radio communication system are described. By adjusting a transmission envelope of a mobile station which transmits information in a burst fashion, harmonics which would normally be introduced into a hearing aid placed in close proximity with the mobile station are avoided. The modified transmission envelope provides a more gradual up and down-ramping of power to the transmitter to avoid introducing harmonics in the audio frequencies. This service can be provided at a system level in a manner which will avoid unnecessarily reducing system capacity by, for example, authenticating a user's identity, his or her subscription to this mode of operation, the mobile station's ability to modify the transmission envelope according to the present invention, and the user's desire to actually implement the hearing aid compatible mode of operation.

FOR THE PURPOSES OF INFORMATION ONLY

Codes used to identify States party to the PCT on the front pages of pamphlets publishing international applications under the PCT.

AM	Armenia	GB	United Kingdom	MW	Malawi
AT	Austria	GE	Georgia	MX	Mexico
AU	Australia	GN	Guinea	NE	Niger
BB	Barbados	GR	Greece	NL	Netherlands
BE	Belgium	HU	Hungary	NO	Norway
BF	Burkina Faso	IE	Ireland	NZ	New Zealand
BG	Bulgaria	IT	Italy	PL	Poland
BJ	Benin	JP	Japan	PT	Portugal
BR	Brazil	KE	Kenya	RO	Romania
BY	Belarus	KG	Kyrgyzstan	RU	Russian Federation
CA	Canada	KP	Democratic People's Republic of Korea	SD	Sudan
CF	Central African Republic	KR	Republic of Korea	SE	Sweden
CG	Congo	KZ	Kazakhstan	SG	Singapore
CH	Switzerland	LI	Liechtenstein	SI	Slovenia
CI	Côte d'Ivoire	LK	Sri Lanka	SK	Slovakia
CM	Cameroon	LR	Liberia	SN	Senegal
CN	China	LT	Lithuania	SZ	Swaziland
CS	Czechoslovakia	LU	Luxembourg	TD	Chad
CZ	Czech Republic	LV	Latvia	TG	Togo
DE	Germany	MC	Monaco	TJ	Tajikistan
DK	Denmark	MD	Republic of Moldova	TT	Trinidad and Tobago
EE	Estonia	MG	Madagascar	UA	Ukraine
ES	Spain	ML	Mali	UG	Uganda
FI	Finland	MN	Mongolia	US	United States of America
FR	France	MR	Mauritania	UZ	Uzbekistan
GA	Gabon			VN	Viet Nam

**CORDLESS TIME-DUPLEX PHONE WITH
IMPROVED HEARING-AID COMPATIBLE MODE**

BACKGROUND

The present invention provides, for example, a method to reduce hum in a
5 hearing aid due to RF electromagnetic compatibility (EMC) effects when a hearing
aid wearer uses a cordless (e.g., cellular) phone employing intermittent
transmission.

Conventional radiocommunication systems provide examples of three
different types of radio transmission protocol for use by cellular or cordless phones.
10 The earlier deployed systems, such as the U.S. AMPS system, the British TACS
system and the Scandinavian NMT system, employ continuous analog FM
transmission of speech, and each conversation is allocated a different channel. This
technique is known as Frequency Division Multiple Access or FDMA.

More recently, digitally coded speech has been employed in systems such as
15 the European GSM system, the U.S. IS-54 system and the Japanese PDC system,
because it affords better protection against fading in the mobile radio propagation
environment, higher system capacity, longer battery life and privacy through digital
encryption. These systems employ Time Division Multiple Access (TDMA)
techniques whereby a single radio frequency channel is shared between several calls
20 by allocating different timeslots to different calls. Additionally, the transmit and
receive timeslots in these TDMA systems are staggered so that the phone does not
have to transmit and receive simultaneously. This feature is advantageous in
eliminating the large, expensive and lossy duplexing filters that were used with the
earlier continuous analog transmission systems that required simultaneous
25 transmission and reception. Moreover, time duplex is essential when the frequency
band allocated for a system does not allow an adequate frequency spacing between
transmit and receive, as in the so-called U.S. "part 15, unlicensed bands", the bands
used for the European digital cordless phone standard known as DECT, and in the
1900MHz PCS bands just released in the USA.

In time duplex operation, the transmitter is inactive for a certain period during each frame, which period is at least long enough to receive a signal burst.

The loss of transmission time caused by this inactive period is made up by buffering the digitized speech signal to be transmitted in a memory and subsequently

5 transmitting the buffered signal at a higher rate than that at which it was buffered during its allocated slot. The peak transmitter power is increased by the same factor to support the higher rate, maintaining the same (or lower) mean power. When such a phone is operated near sound reproducing equipment that is not designed for operation in close proximity to radio transmitters, such as hearing aids or high-fi
10 equipment, the transmitter pulse repetition rate can be spuriously detected in the equipment giving rise to an audible buzz. This effect only occurs when the cordless phone is extremely close to such equipment, for example, within three feet or less, which is the case when a phone is placed next to the hearing aid of a hearing impaired subscriber.

15 A third access method that has been proposed is Code Division Multiple Access (CDMA). CDMA systems may employ continuous transmission but still permit many conversations to take place using the same spectrum, as the differently coded signals can overlap without interference providing their relative power levels are not too divergent. A disadvantage of CDMA phones is their lack of
20 retrospective compatibility with earlier standards, an important factor in introducing new systems, or the larger size and higher cost of providing a dual mode phone with retrospective compatibility. CDMA systems (such as that described in the US IS-95 standard) are also risky precisely because of their need for very strict power control since one errant phone that incorrectly controls its power can jam the whole system.
25 A touted potential advantage of CDMA systems generally is that the phone transmits a lower power continuously instead of using higher power bursts, however the U.S. IS-95 system employs burst transmission of varying duty factor in the phone and thus does not offer much improved hearing aid compatibility.

Hybrid time duplex CDMA systems have been developed to overcome the
30 disadvantages of IS-95, such as the Omnipoint wideband CDMA system developed

for use in the so-called "unlicensed" bands that do not have a frequency duplex spacing. Time-duplex CDMA phones also employ intermittent or burst transmission and so require a method to improve compatibility with the use of hearing aids.

Since the above-mentioned TDMA systems account for over 99% of the
5 current world market, it is important to provide phones which also include an operating mode of improved hearing aid compatibility so that hearing aid wearers are not subjected to the above-described annoying hum or buzz.

SUMMARY

Cordless handheld telephones employing time-duplex transmission are known
10 which advantageously do in not require simultaneous transmission and reception, allowing costs to be reduced and giving longer battery life. The transmitter normally operates for a short time slot in a repetitive frame period to transmit speech to a base station alternating with the receiver operating in a different short time slot in the repetitive frame period to receive speech from the base station.
15 Other telephone calls may be supported simultaneously using the same radio channel by assigning different transmit and different receive slots to different phones.

When a cordless phone is used by a hearing aid wearer, the phone comes in close proximity to the hearing aid and, in particular, the cordless phone's transmit antenna is only inches from the hearing aid, potentially giving rise to hum due to
20 the hearing aid spuriously detecting the transmitter's RF pulse repetition frequency. Phones according to the present invention include both the normal mode of operation and a new mode of operation which can be invoked by hearing aid wearers when the service provider has issued a proper authorization with the subscription. The subscription offered to hearing aid users can also include other
25 features useful to the hearing impaired such as text message services.

In the hearing aid compatible mode, phones according to exemplary embodiments of the present invention adopt a sinusoidal transmission amplitude envelope in which annoying high harmonic frequencies of the transmission pulse are suppressed. This technique causes a user's transmission to occupy a larger

proportion of the repetitive frame period then that used in the normal mode of operation, thereby reducing the number of users that can be accommodated on the channel. However, if only authorized hearing aid users are eligible to transmit in the new mode, the loss of capacity to the system overall will be negligible. The present invention provides an improved pulse shaping for intermittent transmitters using any access methodology, e.g., TDMA, CDMA or time-duplex CDMA, to eliminate pulse harmonics that cause the buzzing anomaly in sound reproducing equipment mentioned above.

BRIEF DESCRIPTION OF THE DRAWINGS

The foregoing, and other, objects, features and advantages of the present invention will become more apparent upon reading the following detailed description in conjunction with the drawings in which:

Figure 1 illustrates an example of the transmission format used in the GSM system, which format is used as an example to describe the present invention;

Figure 2 illustrates an exemplary transmission format for the GSM system modified according to the present invention;

Figure 3 is a block diagram illustrating an exemplary system according to the present invention; and

Figure 4 is a flowchart used to illustrate an exemplary method according to the present invention.

DETAILED DESCRIPTION

Figure 1 shows an example of the transmission format employed in the TDMA cellular system known as Global System for Mobile Communications (GSM). Therein, a repetitive frame period of approximately 4.6mS is divided into eight timeslots (or 16 timeslots in the half-rate mode). Mobiles receive on a radio channel in a receive frequency band, each mobile receiving in an allocated one of the eight timeslots, e.g., timeslot 1 in Figure 1. The mobile also transmits on a channel in the transmit frequency band in an allocated transmit timeslot, e.g.,

timeslot 1. Transmit timeslots are staggered in time relative to their corresponding receive timeslots so that the mobile phone does not have to transmit and receive at the same time. This is illustrated in the example of Figure 1 as the time duplex spacing of two timeslots between timeslot 1 on the mobile receive frequency and timeslot 1 on the mobile transmit frequency. When different transmit and receive bands are employed, the base station may receive and transmit in all timeslots, and thus support eight conversations in the same transmit/receive channel pair. Otherwise, if no frequency-duplex spacing is employed, the base station can, for example, use four of the timeslots for transmission and four for reception, thus supporting four conversations in a single transmit/receive channel. In GSM, the remainder of the time not used by a particular mobile for transmission and reception is available for the mobile's receiver to scan other channels in order to determine, for example, when it should request a handover to stronger base station. This process is known as mobile-assisted handover.

The transmit pulse formats are defined by the GSM specifications and include a guard time between pulses, an up/down-ramping time to turn the transmitter power smoothly on and off to avoid spectral splatter into unsynchronized adjacent channels, some guard bits at each end of the burst, and the main body of the TDMA burst which transmits 114 data bits, 26 sync bits and two control bits at a flat power level. The up/down-ramping times are adequate to prevent spectral spreading into an adjacent channel 200KHz away, but are not designed to control harmonics of the 220Hz pulse repetition rate in the audio band of 300Hz to 4KHz. Absent the special transmission mode described below, these harmonics cause audible noises in a hearing impaired user's hearing aid.

Figure 2 shows an exemplary modified GSM reception and transmission format designed to control harmonics of the frame frequency. The receive slots are the same as those illustrated in Figure 1, but the transmit slot timing has been centered midway between receive slots. This permits a relatively slow, almost sinusoidal, up and down-ramping of the transmitter power. The central part of the peak (e.g., transmit timeslot 2) can be used for transmitting a normal burst format,

e.g., comprising the above-described data and sync bits. Optionally, one or more of the slots on either side of the center slot may be used to repeat transmission of the data, allowing a special base station receiver to utilize the extra energy to improve signal quality and potentially allow the peak power to be reduced. In this exemplary embodiment, the transmitter power is held to zero or very low for approximately one timeslot (e.g., 0.56ms) after the end of reception (e.g., transmit slot 7 in Figure 2) and before the start of the next receive burst (e.g., transmit timeslot 5), to allow settling time for frequency synthesizer frequency changes between transmit and receive, so that the transmitter is not generating RF energy until the frequency is at least within the allotted 200KHz channel.

Using the exemplary format illustrated in Figure 2, there are essentially three slots per frame of more or less total transmitter silence in which other mobile transmitters can be received interference free. Thus in total, the capacity is reduced by a factor of two, for the fraction of a percent of cases where a hearing impaired user needs to employ the inventive mode. The impact on system capacity is therefore virtually negligible if use is restricted to authorized subscribers. Having described an exemplary mobile transmission mode for hearing-impaired subscribers, an exemplary system level implementation will now be described.

Figure 3 represents a block diagram of an exemplary cellular mobile radiotelephone system according to one embodiment of the present invention which can be used to implement the foregoing. The system shows an exemplary base station 110 and a mobile 120. The base station includes a control and processing unit 130 which is connected to the MSC 140 which in turn is connected to the public switched telephone network (not shown).

The base station 110 for a cell includes a plurality of voice channels handled by voice channel transceiver 150 which is controlled by the control and processing unit 130. Also, each base station includes a control channel transceiver 160 which may be capable of handling more than one control channel. The control channel transceiver 160 is controlled by the control and processing unit 130. The control channel transceiver 160 broadcasts control information over the control channel of

the base station or cell to mobiles locked to that control channel.

When the mobile 120 first enters the idle mode, it periodically scans the control channels of base stations like base station 110 to determine which cell to lock to on or camp on. The mobile 120 receives the absolute and relative information broadcast on a control channel at its voice and control channel transceiver 170. Then, the processing unit 180 evaluates the received control channel information which includes the characteristics of the candidate cells and determines which cell the mobile should lock to. The received control channel information not only includes absolute information concerning the cell with which it is associated, but also contains relative information concerning other cells proximate to the cell with which the control channel is associated. These adjacent cells are periodically scanned while monitoring the primary control channel to determine if there is a more suitable candidate. Analogously, this equipment can be used to perform the MAHO techniques described previously while the mobile 120 is connected to the system. Additional information relating to specifics of mobile and base station implementations can be found in copending U.S. Patent Application Serial No. 07/967,027 entitled "Multi-Mode Signal Processing" filed on October 27, 1992 to P.Dent and B. Ekelund, which disclosure is incorporated here by reference.

In the context of the present invention, processing unit 180 of mobile 120 controls the tuning and power used by the transceiver 170. Thus, when operating in a normal mode, processing unit 180 will control transceiver 170 in, for example, the manner illustrated in Figure 1. When, on the other hand, operating in a hearing impaired mode, processing unit 180 will more gradually rampup and rampdown the power used in transceiver 170. Although one exemplary power profile is illustrated in Figure 2, those skilled in the art will appreciate that the more general concept of the present invention, i.e., the gradual ramping of power up and down for burst transmissions in a second operational mode, need not occupy five timeslots as illustrated in Figure 2. Depending upon the transmission format and system in which the present application is applied, and the system designers desire to

accommodate hearing impaired users while also not impairing capacity should the number of hearing impaired users be relatively large, e.g., in a specific microcell environment, the number of timeslots over which a single burst can be power tapered may be more or fewer than five. As a more general formulation, it is
5 sufficient to say that power ramping according to the present invention will simply occupy more than one timeslot in a TDMA type format.

The GSM system includes a cipher-protected authentication system to verify the identity of users. This is embodied in a plug-in smart card (not shown) disposed in the mobile unit which contains a subscriber-specific secret key known only to the
10 service-provider's computer and stored in the Home Location Register (HLR). When a mobile phone accesses a network station, the network station verifies the identity of the user as determined by the installed smart card by contacting the HLR and obtaining a set of three security numbers. The first number is a random challenge which is sent to the mobile station and passed to the installed smart card.
15 The smart card combines the challenge with the secret key and outputs a response dependent on both. Part of the response is sent by the mobile station to the network station, where the network station compares it with the second security number received from the HLR. If the response matches the second number, then the mobile station is verified to have a valid smart card installed which contains the
20 secret key issued only to that subscriber. The other part of the smart card's response is used as a cipher key for encrypting the rest of the call, and corresponds to the third security number sent from the HLR to the network station. Only if this cipher key also matches can communication actually take place. The GSM authentication and cipher system thus guarantee the identity of the subscriber. The
25 HLR can then also access other subscriber-related information such as whether international calls are permitted or not, whether the hearing-impaired mode is authorized or not, and so-on.

With reference to the flowchart of Figure 4, the network station can also request the mobile phone to transmit its electronic serial number by transmitting an
30 interrogation signal thereto at block 200. A data base of electronic serial numbers

may be maintained by the network to look-up characteristics of each equipment, such as whether the hearing-impaired mode is supported by the equipment. There are also "class-bits" stored in the mobile which can be used to indicate the type of equipment to the network. In either case, the system determines the mobile's capabilities as indicated by block 210. Additionally, the user can program the phone from the keyboard and display to enable the hearing impaired mode automatically whenever a call is placed or received, or selectively as desired. This information can be transmitted to the network at, for example, registration. The network can check all three types of information, namely:

- 10 (i) that the hearing impaired mode is an allowed mode in the subscriber profile (in which case it can be offered cost free) at block 220;
- (ii) that the phone currently in use supports the hearing impaired (HI) mode (block 230),
- and
- 15 (iii) that the user has enabled the hearing impaired mode (block 240).

According to the exemplary embodiment illustrated in Figure 4, all three conditions should be satisfied in order for the hearing impaired mode, e.g., the transmission format of Figure 2, to be allocated by, for example, setting an indicating bit in the channel-allocation message at call set up at block 250. If one of these conditions is not met, then the flow proceeds to block 260 where normal mobile transmission (e.g., Fig. 1) is performed by, for example, not setting the indicating bit.

Alternatively, if a user does not have the hearing impaired mode enabled in his or her subscriber profile this user can still receive the mode upon request if the equipment supports it, but the operator may then charge a premium (for example, double airtime costs) for the extra system capacity consumed. This extra charge can be waived, at the operator's discretion, for hearing impaired users that register to use the mode.

A method has been described for improving hearing aid compatibility in digital cellular phone systems. Other solutions that improve service to the hearing

impaired may, like the solution described above, consume more system capacity when employed thus reducing operator revenues. for example, in a system such as IS-95, the varying duty factor used to reduce average system loading consumed by one telephone call may be omitted to improve hearing aid compatibility. In a

5 TDMA system, it is possible to conceive of allocating all timeslots to one user if a frequency duplexer is used, thus eliminating on-off transmitter switching. The reduction in system capacity and revenue is however insignificant when use is restricted to the small, genuine hearing impaired community however, which is guaranteed by the automatic authentication aspect of the invention. Any such system

10 as may be devised or modified by persons skilled in the art making use of the above principles is deemed to fall within the scope of this invention as described by the following claims.

CLAIMS:

1. In a cellular telephone network including a network of base stations for communicating with a plurality of mobile stations, a method of providing service to a hearing impaired user comprising the following steps:
 - 5 transmitting an interrogation signal from the network to a mobile station and receiving a response verifying an identity of the user; and
 - allocating a higher proportion of system communications resources if requested by the user and if a profile contained in a network database for said identified user permits such allocation.
- 10 2. The method of claim 1 in which the step of allocating said higher proportion of system resources includes the step of allocating one or more additional timeslots to the mobile station's transmitter.
3. The method of claim 1 in which the step of allocating said higher proportion of system resources includes the step of allowing said mobile station to
15 increase its average transmit duty factor.
4. The method of claim 2 further comprising the step of reducing a peak power associated with said mobile station's transmitter in relation to the one or more additional timeslots allocated.
5. The method of claim 3 further comprising the step of reducing a peak
20 power associated with said mobile station's transmitter in relation to the increase in average transmit duty factor.
6. The method of claim 2 in which said mobile transmitter employs said additional timeslots to extend power up- and down-ramping time to reduce audio frequency harmonics of a pulse repetition rate associated with said mobile station's
25 transmitter.

7. In a cellular telephone network including a network of base stations for communicating with a plurality of mobile telephones, a method of improving service to the hearing impaired user comprising the following steps;

transmitting an interrogation signal from the network to a mobile
5 station to determine capabilities of said mobile station and verify a user identity;
receiving a response from the mobile station authenticating the user identity,
indicating the capabilities of said mobile station and including a request for special
user features; and
allocating a hearing aid compatible communications channel if indicated as
10 said special user request and if said requested feature is supported by said indicated
capabilities.

8. The method of claim 7 in which said step of allocating further comprises the step of:

allocating said hearing aid compatible channel only if said identified
15 user is registered as authorized to receive said hearing aid compatible channel.

9. The method of claim 7 in which said step of allocating further comprises the step of:

allocating said hearing aid compatible channel with no additional
airtime cost being accumulated by a network billing system if said identified user
20 has previously registered to receive said hearing aid compatible channel.

10. The method of claim 6 in which said power up- and down-ramping is an approximately sinusoidal amplitude modulation.

11. A mobile station comprising:

a transceiver for transmitting bursts of information at controllable
25 power levels; and
a processor for controlling power levels associated with said bursts

13

according to one of a first mode and a second mode.

12. The mobile station of claim 11 wherein said first mode is a normal operating mode and said second mode is a hearing aid compatible operating mode.

5 13. The mobile station of claim 12 wherein said processor adjusts said power levels in said second mode at a rate which is substantially more gradual than a rate associated with said first mode.

14. The mobile station of claim 12 wherein said processor adjusts said power levels in said second mode to mitigate harmonics which can be induced in a hearing aid by said mobile station.

10 15. The mobile station of claim 11 wherein said transceiver receives an indicating bit at call set-up and said processor controls said transceiver in accordance with one of said first and second modes based upon a value of said indicating bit.

FIG. 1

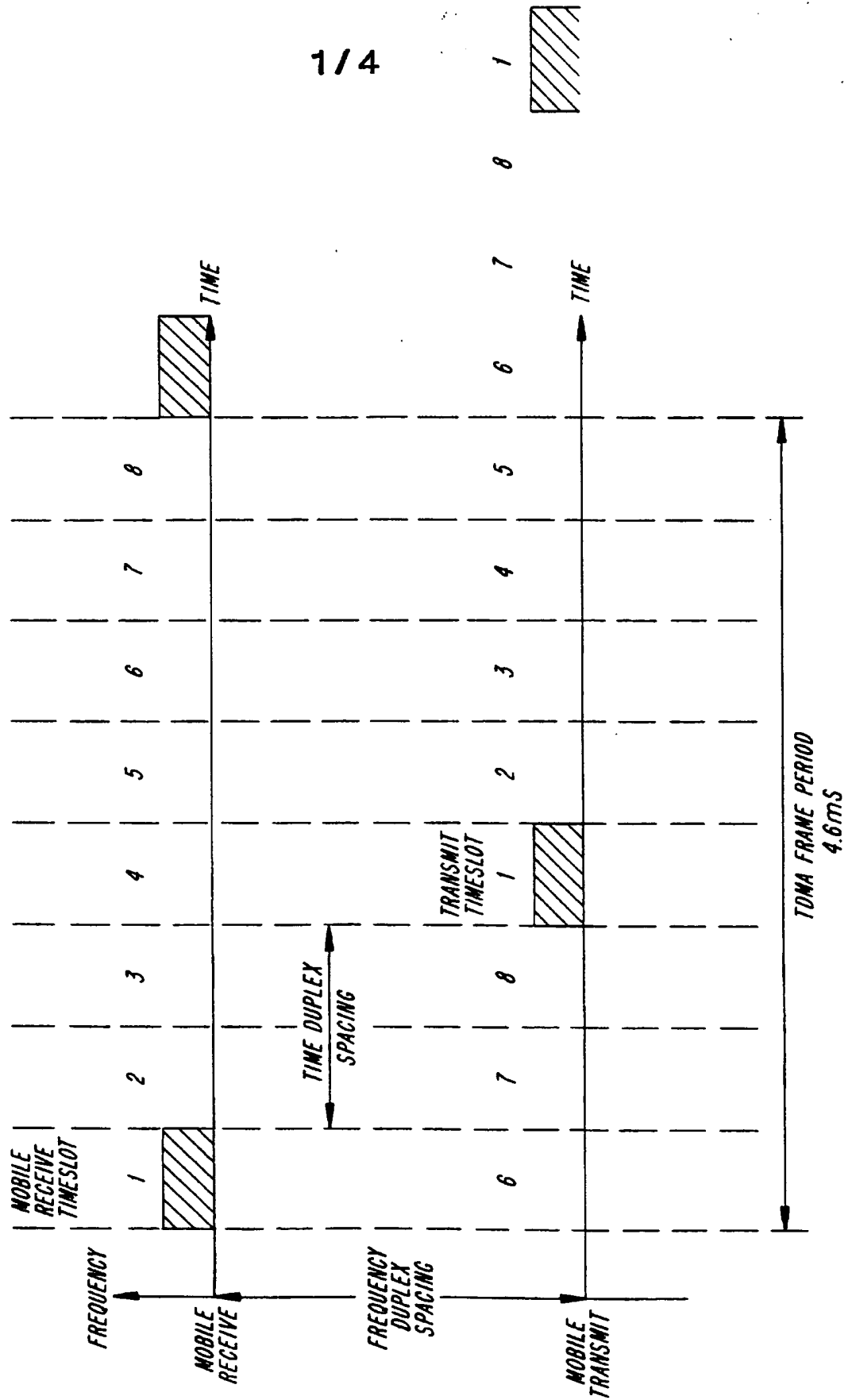
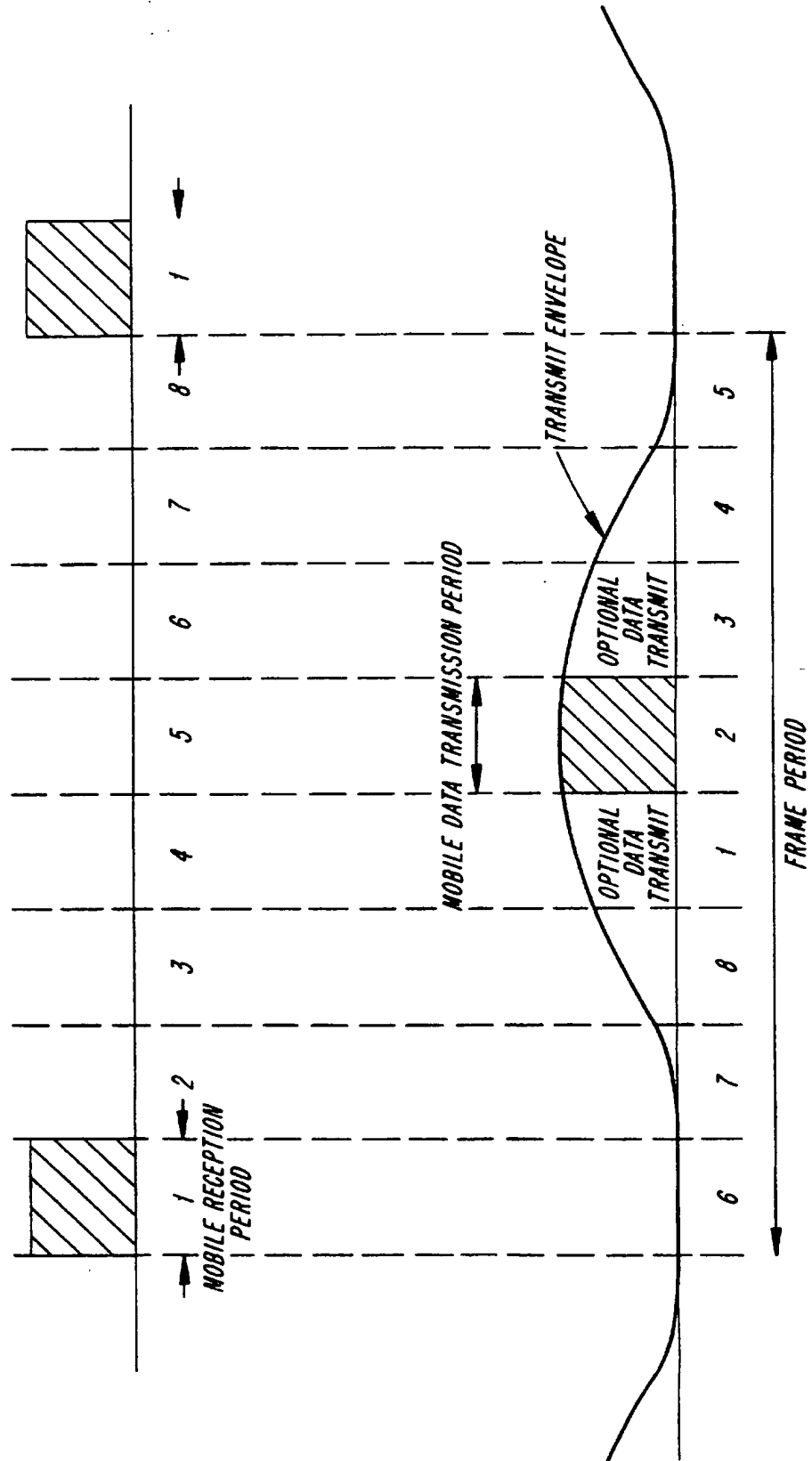
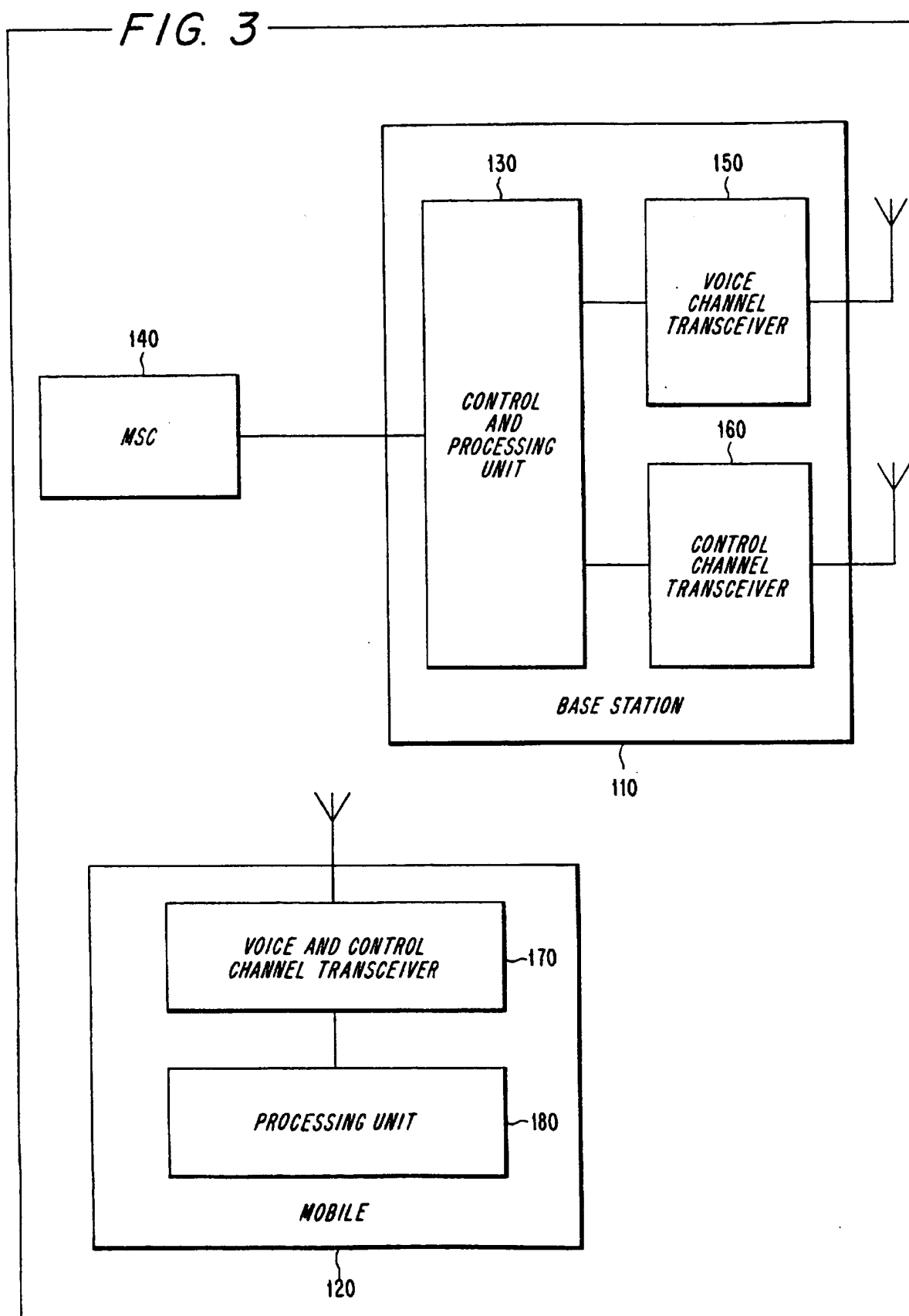


FIG. 2

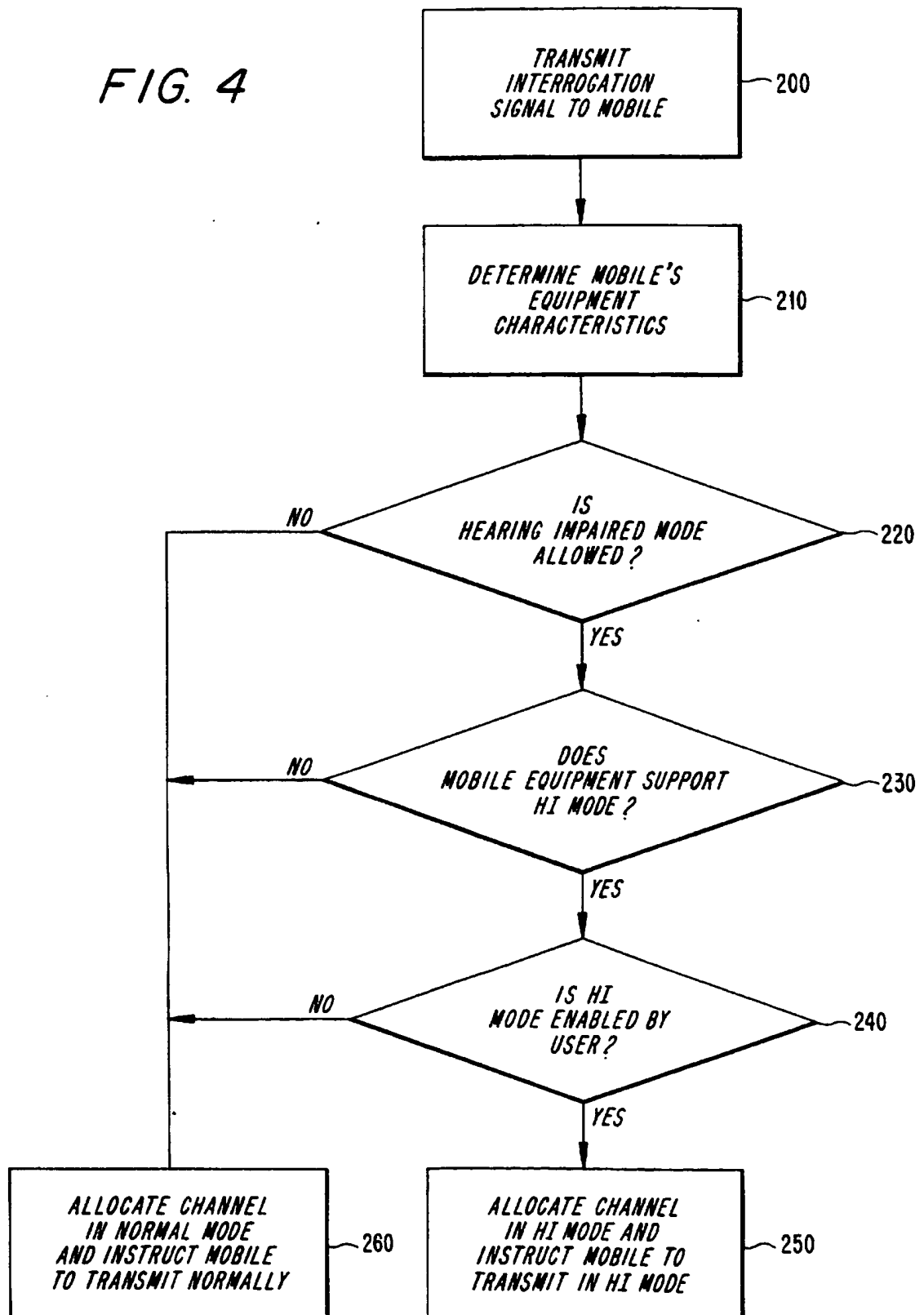


3/4



4 / 4

FIG. 4



INTERNATIONAL SEARCH REPORT

International Application No
PCT/US 97/00711

A. CLASSIFICATION OF SUBJECT MATTER
IPC 6 H04Q7/22 H04B15/00

According to International Patent Classification (IPC) or to both national classification and IPC

B. FIELDS SEARCHED

Minimum documentation searched (classification system followed by classification symbols)

IPC 6 H04Q H04B

Documentation searched other than minimum documentation to the extent that such documents are included in the fields searched

Electronic data base consulted during the international search (name of data base and, where practical, search terms used)

C. DOCUMENTS CONSIDERED TO BE RELEVANT

Category *	Citation of document, with indication, where appropriate, of the relevant passages	Relevant to claim No.
A	GB 2 238 449 A (TRADE AND INDUSTRY SECRETARY 0) 29 May 1991 see page 1, line 21 - line 29 see page 2, line 20 - page 3, line 20 ---	1,7
A	EP 0 685 972 A (NOKIA MOBILE PHONES LTD) 6 December 1995 see column 1, line 51 - column 2, line 28 ---	1,7
X	EP 0 688 109 A (NOKIA MOBILE PHONES LTD) 20 December 1995	11
A	see column 1, line 49 - column 2, line 31 see column 4, line 29 - line 56 ---	15
-/--		

☒ Further documents are listed in the continuation of box C.

☒ Patent family members are listed in annex.

* Special categories of cited documents :

- *A* document defining the general state of the art which is not considered to be of particular relevance
- *E* earlier document but published on or after the international filing date
- *L* document which may throw doubts on priority claim(s) or which is cited to establish the publication date of another citation or other special reason (as specified)
- *O* document referring to an oral disclosure, use, exhibition or other means
- *P* document published prior to the international filing date but later than the priority date claimed

- *T* later document published after the international filing date or priority date and not in conflict with the application but cited to understand the principle or theory underlying the invention
- *X* document of particular relevance; the claimed invention cannot be considered novel or cannot be considered to involve an inventive step when the document is taken alone
- *Y* document of particular relevance; the claimed invention cannot be considered to involve an inventive step when the document is combined with one or more other such documents, such combination being obvious to a person skilled in the art.
- *&* document member of the same patent family

Date of the actual completion of the international search

23 June 1997

Date of mailing of the international search report

- 4. 07. 97

Name and mailing address of the ISA

European Patent Office, P.B. 5818 Patentlaan 2
NL - 2280 HV Rijswijk
Tel. (+ 31-70) 340-2040, Tx. 31 651 epo nl,
Fax (+ 31-70) 340-3016

Authorized officer

Gerling, J.C.J.

INTERNATIONAL SEARCH REPORT

International Application No
PCT/US 97/00711

C.(Continuation) DOCUMENTS CONSIDERED TO BE RELEVANT

Category *	Citation of document, with indication, where appropriate, of the relevant passages	Relevant to claim No.
X	EP 0 632 609 A (AT & T CORP) 4 January 1995	11
A	see column 1, line 20 - line 35 see column 2, line 20 - line 26 see column 2, line 32 - line 44 ---	1,7
A	FUNKSCHAU, vol. 67, no. 25, 24 November 1995, page 60/61 XP000548879 "BRUMMEN IM OHR-GSM-HANDYS UN HOERGERAETE" -----	

INTERNATIONAL SEARCH REPORT

Information on patent family members

International Application No

PCT/US 97/00711

Patent document cited in search report	Publication date	Patent family member(s)	Publication date
GB 2238449 A	29-05-91	NONE	
EP 0685972 A	06-12-95	FI 942548 A	01-12-95
EP 0688109 A	20-12-95	FI 942839 A	16-12-95
		JP 8084113 A	26-03-96
EP 0632609 A	04-01-95	CN 1103220 A	31-05-95
		JP 7154353 A	16-06-95